

What is claimed is  
~~U.S. CLAIMS~~

1. A method of designing a digital filter, including the steps of  
determining a real-valued discrete-frequency representation of a desired  
full length digital filter;  
transforming said discrete-frequency representation into a correspond-  
ing discrete-time representation;  
circularly shifting said discrete-time representation; and  
applying a shortening window to said discrete-time representation to  
produce a zero-padded reduced length filter.
2. The method of claim 1, <sup>further</sup> including the step of circularly shifting said  
reduced length filter to remove leading zeroes.
3. The method of claim 1 or 2, wherein said real-valued discrete-frequency  
representation is formed by a noise suppressing spectral subtraction algo-  
rithm.
4. The method of claim 1 or 2, wherein said real-valued discrete-frequency  
representation is formed by a frequency selective non-linear algorithm for echo  
cancellation.
5. The method of claim 1, wherein said window is a Kaiser window.
6. The method of claim 1, <sup>further</sup> including the step of transforming said reduced  
length filter into a minimum phase filter.
7. A digital convolution method, including the steps of  
determining a real-valued discrete-frequency representation of a desired  
full length digital filter;  
transforming said discrete-frequency representation into a correspond-  
ing discrete-time representation;  
circularly shifting said discrete-time representation;

applying a shortening window to said discrete-time representation to produce a zero-padded reduced length filter; and

convolving an input signal with said zero-padded reduced length filter.

5 8. The method of claim 7, <sup>Further</sup> including the step of circularly shifting said reduced length filter to remove leading zeroes.

9. The method of claims 7, <sup>Further</sup> including the step of transforming said reduced length filter into a minimum phase filter.

10. The method of claim 7, <sup>Further</sup> 8 or 9, including the step of performing the convolution step in the time domain using the discrete-time representation of said reduced length filter.

11. The method of claim 7, <sup>Further</sup> 8 or 9, <sup>and</sup> including the step of performing the convolution step in the frequency domain by using the overlap-add method.

12. The method of claim 7, 8 or 9, including the step of performing the convolution step in the frequency domain by using the overlap-save method.

13. A digital filter design apparatus, including  
means for determining a real-valued discrete-frequency representation of a desired full length digital filter;

means for transforming said discrete-frequency representation into a corresponding discrete-time representation;

means for circularly shifting said discrete-time representation; and

means for applying a shortening window to said discrete-time representation to produce a zero-padded reduced length filter.

14. The apparatus of claim 13, <sup>Further</sup> including means for circularly shifting said reduced length filter to remove leading zeroes.

✓ 15. The apparatus of claim 13 ~~or 14~~, wherein said window applying means implements a Kaiser window.

✓ 16. The apparatus of claim 13, <sup>Further</sup> including means for transforming said reduced length filter into a minimum phase filter.

17. A digital convolution apparatus, including  
means for determining a real-valued discrete-frequency representation of a desired full length digital filter;

means for transforming said discrete-frequency representation into a corresponding discrete-time representation;

means for circularly shifting said discrete-time representation;

means for applying a shortening window to said discrete-time representation to produce a zero-padded reduced length filter; and

means for convolving an input signal with said zero-padded reduced length filter.

18. The apparatus of claim 17, <sup>Further</sup> including means for circularly shifting said reduced length filter to remove leading zeroes.

✓ 19. The apparatus of claims 17, <sup>Further</sup> including means for transforming said reduced length filter into a minimum phase filter.

✓ 20. The apparatus of claim 17, ~~18 or 19~~, <sup>Further</sup> including means for performing the convolution step in the time domain using the discrete-time representation of said reduced length filter.

✓ 21. The apparatus of claim 17, ~~18 or 19~~, <sup>Further</sup> including means for performing the convolution step in the frequency domain by using the overlap-add method.

✓ 22. The method of claim 17, ~~18 or 19~~, <sup>Further</sup> including means for performing the convolution step in the frequency domain by using the overlap-save method.